

USSN 10/042,237

Art Unit 2644

**Amendments to Specification**

Please correct numbered paragraph [0015] as follows:

- a1*  
— [0015] The invention provides a correlation-based matrix is-generated using zero-lag auto and cross-correlations of signals commonly found in echo cancellers. --

Please amend numbered paragraph [0020] as follows

- a2*  
— [0020] The invention will now be described in more detail, by way of example only, with reference to the accompanying drawings, in which:-

Figure 1 is a schematic diagram of an echo canceller using LMS Adaptive Filtering; and

Figure 2a is a plot showing the value of det [R] under normal convergence;

Figure 2b is a plot showing the value of det [R] with a path change; and

Figure 2c is a plot showing the value of det [R] with double-talk; and

Figure 3 illustrates the process of detecting double-talk. --

Please replace the numbered paragraphs [0023], [0024], [0029], [0030], [0031], [0033], [0037], [0038], [0039] with the following new paragraphs as follows. In each case the superfluous  $\mu$  has been deleted:

- [0023] The preferred embodiment of the algorithm for this patent uses the Normalized-LMS (N-LMS) algorithm. Mathematically, the adaptive filter tap-weight update procedure for the N-LMS algorithm consists of the following three equations

$$\hat{d}[n] = \hat{w}^H[n]u[n]$$

$$e[n] = d[n] - \hat{d}[n]$$

$$\hat{w}[n+1] = \hat{w}[n] + \frac{\mu}{a + \|u[n]\|^2} u[n]e[n]$$

*a3*  
— [0024] where

$u[n] = R_{IN}$  = echo source signal

$w[n]$  = LMS filter coefficients

*a3*  
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$d[n] = S_{IN}$  = desired LMS output (echo + double-talk)

$\hat{d}[n] =$  LMS output (estimated echo)

$e[n] = S_{OUT}$  = LMS error signal

$\mu$  = LMS step-size parameter

$a$  = A small constant (provides numerical stability). —

— [0029] Consider two signals,  $X_0[n]$  and  $X_1[n]$  generated by a linear combination of two real-valued source signals,  $S_0[n]$  and  $S_1[n]$ . Mathematically, this mixing process may be described as

$$X_i = H_{i,0} \cdot S_0 + H_{i,1} \cdot S_1, \quad i = 0, 1$$

— [0030] where  $H_{i,j}$  are the mixing coefficients. In matrix form, this may be written as

$$\mathbf{X} = \mathbf{H} \cdot \mathbf{S} \quad \text{---}$$

— [0031] where

$$\mathbf{X} = \begin{bmatrix} X_0 \\ X_1 \end{bmatrix}, \mathbf{H} = \begin{bmatrix} H_{0,0} & H_{0,1} \\ H_{1,0} & H_{1,1} \end{bmatrix} \text{ and } \mathbf{S} = \begin{bmatrix} S_0 \\ S_1 \end{bmatrix} \quad \text{---}$$

— [0032] A matrix  $\mathbf{R}$  is defined as

$$\mathbf{R} = E[\mathbf{X}\mathbf{X}^T] \quad \text{---}$$

— [0033] where  $E[\dots]$  is the statistical expectation operator.  $\mathbf{R}$  may be expanded in two ways

$$\mathbf{R} = E \begin{bmatrix} X_0 X_0^T & X_0 X_1^T \\ X_1 X_0^T & X_1 X_1^T \end{bmatrix}$$

$$= E[\mathbf{H}\mathbf{S}\mathbf{S}^T\mathbf{H}^T] \quad \text{---}$$

— [0036] The way in which the matrix can be used to perform double-talk and path change detection will now be explained. First, suppose we generate the signal mixtures in using convolutions:

$$\mathbf{X} = \mathbf{H} \otimes \mathbf{S} \quad \text{---}$$

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-- [0037] Now the terms in the mixing matrix can be vectors. We further impose the condition that  $H$  have the following form:

$$H = \begin{bmatrix} H_{0,0} & 1 \\ H_{1,0} & 0 \end{bmatrix} --$$

-- [0038] With  $H$  defined in this way, it is now possible to connect the terms in the preceding equations with the parameters available in the echo canceller layout shown in Figure 1. Let

*Ans*  
*cont*

$S_0$  = echo source signal =  $R_{IN} = u[n]$

$S_1$  = doub c-talk signal

$H_{0,0}$  = echo path

$H_{1,0}$  = LMS filter coefficients =  $\hat{w}[n]$

--[0039] With these definitions, it is apparent that

$$X_0 = H_{0,0} \otimes S_0 + S_1 = S_n = d[n]$$

$$X_1 = H_{1,0} \otimes S_0 = \hat{d}[n]$$

As shown in Figure 3, in practising the invention, a first step 10 is performed to generate the correlation-based matrix R from X<sub>0</sub> and X<sub>1</sub>. A matrix operation 11, for example, forming the determinant is next performed on the determinant, and the result of the matrix operation is then examined at step 12 to detect double-talk and path changes. In the case of the determinant, this is compared with a threshold value. --